

Application of Linux Audio in Hearing Aid Research

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Introduction and Overview

Hearing Aids

Algorithms for Digital Hearing Aids

Evaluation of Hearing Aids

A Linux-based Hearing Aid

RT performance

CPU and battery performance

Delay Constraints

Conclusions

Hearing Aids

Analogue hearing aids

- ▶ Limited capabilities:
- ▶ Frequency shaping and amplification
- ▶ Dynamic compression
- ▶ Static notch filters for feedback cancellation

Hearing Aids

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Digital hearing aids

- ▶ First hearing aid with Digital Signal Processing in 1996
- ▶ New algorithms with no analogue counterpart
- ▶ Wireless binaural link between ears
 - ▶ Parameter exchange
 - ▶ Coming soon: Low-delay low-power audio transmission

Limitations of hearing aids

Device

- ▶ Small battery capacity
- ▶ Low processing power

Acoustics

- ▶ Acoustic feedback
- ▶ Limited audio bandwidth (by tube between receiver and ear)

End user

- ▶ High expenses
- ▶ Unfulfilled expectations

Algorithms scalable to individual hearing loss

- ▶ Dynamic compression
- ▶ Frequency shaping
- ▶ Frequency compression

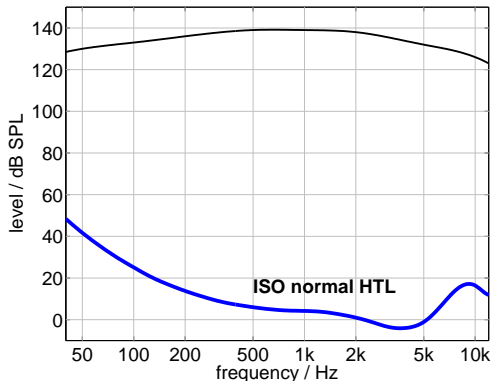
Signal enhancement

- ▶ Directional microphones
- ▶ Noise reduction

Artifact reduction

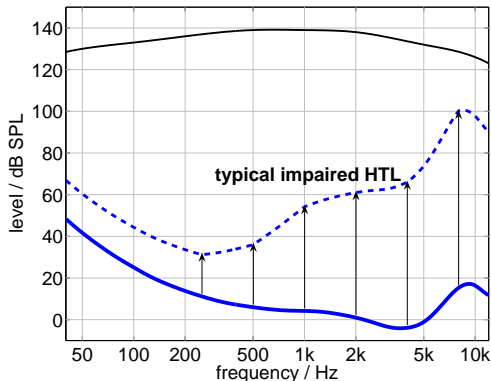
- ▶ Feedback cancellation
- ▶ Future: Non-linearity compensation, ...

Why dynamic compression?



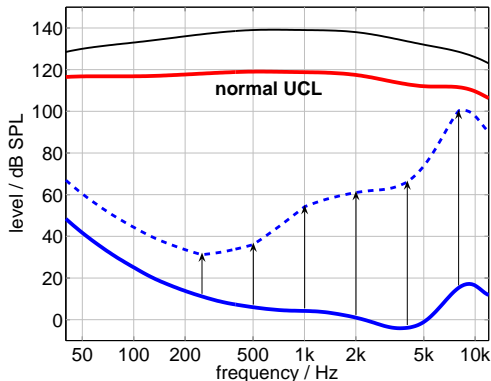
► Normal hearing threshold

Why dynamic compression?



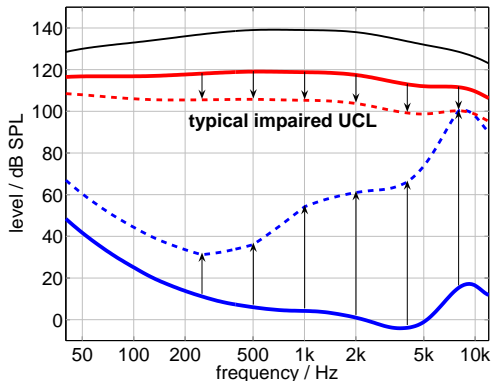
- ▶ Hearing impaired threshold
- ▶ Amplification and frequency shaping

Why dynamic compression?



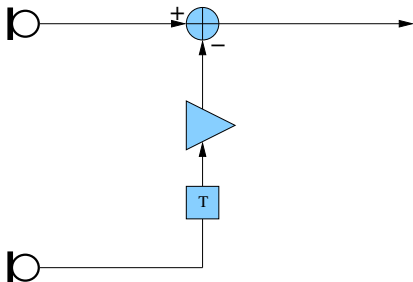
- Normal uncomfortable level

Why dynamic compression?



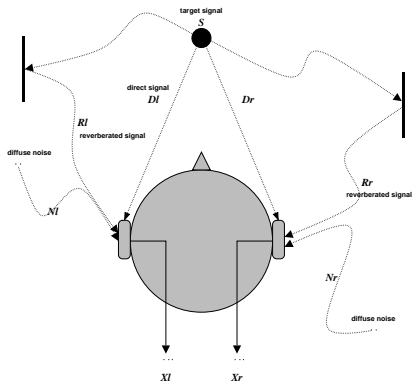
- ▶ Hearing impaired uncomfortable level
- ▶ Dynamic compression

Directional microphones: SNR improvement



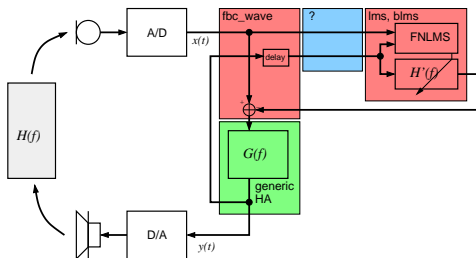
- ▶ Delay-and-sum:
up to 3 dB
SNR-improvement for
on-axis sounds in diffuse
noise
- ▶ Many more beamformers
and direction-of-arrival
estimators

Noise reduction and de-reverberation



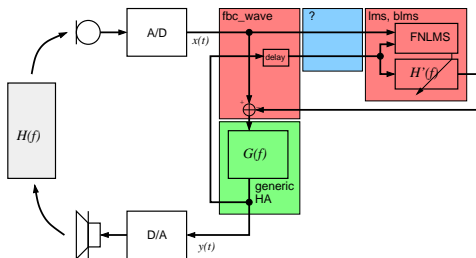
- ▶ Single channel noise reduction
- ▶ Binaural noise reduction, de-reverberation
- ▶ Binaural coherence estimation: Interaural phase difference statistics

Artifact reduction: Feedback cancellation



- ▶ Adaptive feedback cancellation
- ▶ Estimation of feedback signal
- ▶ Subtraction from input

Artifact reduction: Feedback cancellation



- ▶ Adaptive feedback cancellation
- ▶ Estimation of feedback signal
- ▶ Subtraction from input
- ▶ Other solutions: Frequency shifting, phase modulation, binaural coherence

Evaluation of Hearing Aids

Expected benefit of hearing aids

- ▶ Improvement of speech intelligibility
- ▶ Reduction of listening effort in adverse listening conditions
- ▶ Increase of listening comfort

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Evaluation is required!

Evaluation methods

'Objective' methods

- ▶ SNR improvement (e.g., shadow filtering)
- ▶ Speech intelligibility index (f -weighted SNR)
- ▶ Quality prediction (usually similarity measures)

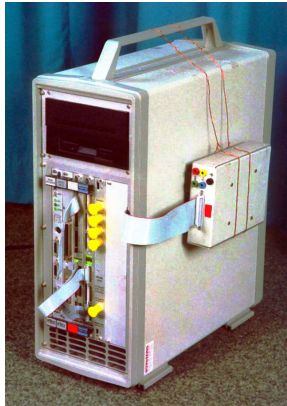
Subjective methods

- ▶ Speech recognition threshold in quiet and noise
- ▶ Quality rating, paired comparison ...

Test conditions

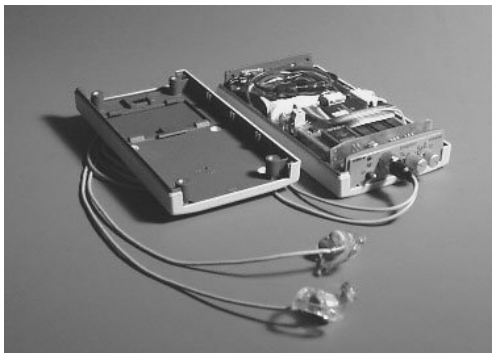
- ▶ 'static' situations:
no spatial influence, pre-processed or real-time processing
- ▶ dynamic conditions:
head movements or moving sources, real-time processing
- ▶ realistic devices, acoustic feedback:
low-delay real-time processing
- ▶ Real-life conditions:
wide dynamic range, portable low-delay real-time processing

Real-time systems



1995: PC with DSP board, assembler programming

Real-time systems



2000: Portable DSP board, assembler programming
(DASi, U. Rass 2001)

Real-time systems



2008: Portable PC, C++ (or Matlab) programming

A Linux Hearing Aid

A Linux Hearing Aid

for research, development and field testing

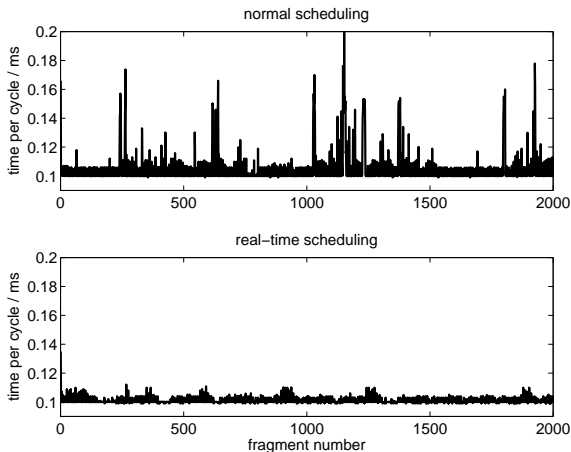
The 'Master Hearing Aid' software

- ▶ Platform for hearing aid algorithm development and evaluation
- ▶ Many audio backends: JACK, ALSA (Linux), ASIO (Windows), file, network, Matlab
- ▶ Extremely modular structure
- ▶ Hearing-aid (and hearing) related processing blocks
- ▶ Commercial product (closed source) by HoerTech gGmbH, Oldenburg
- ▶ Used in research projects and hearing aid industry

Licensing in competitive research

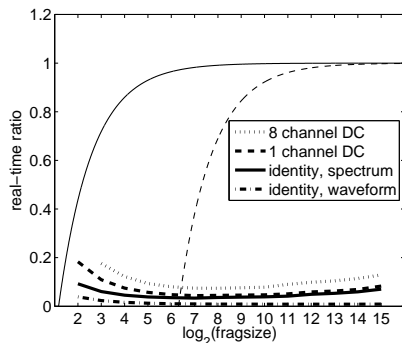
- ▶ Industrial/competitive context often requires closed-source development
- ▶ Industry decision makers fear the open source licensing of Linux
- ▶ It is possible to develop closed software on the Linux platform without infringing any licenses!
- ▶ Careful consideration of what components to use
- ▶ Knowledge of the relevant licenses, and compliance with their terms.
- ▶ Use of open source software in industrial context beneficial for both sides

Comparison of RT kernel versus non-RT kernel



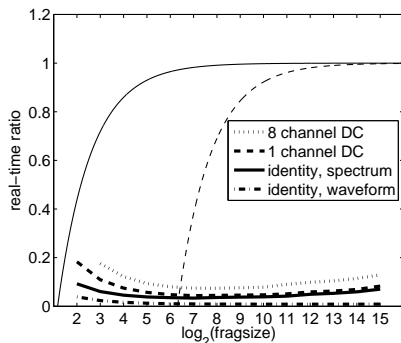
Kernel version 2.6.X (2.6.23?), from Grimm et al. (2006)

RT performance and low-delay

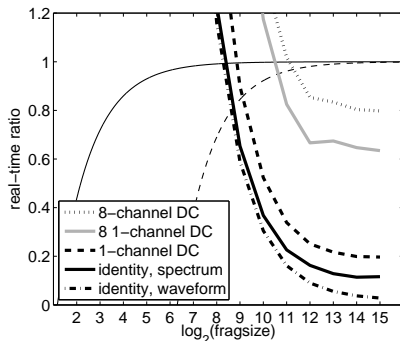


Algorithm implementation in
C/C++

RT performance and low-delay



Algorithm implementation in
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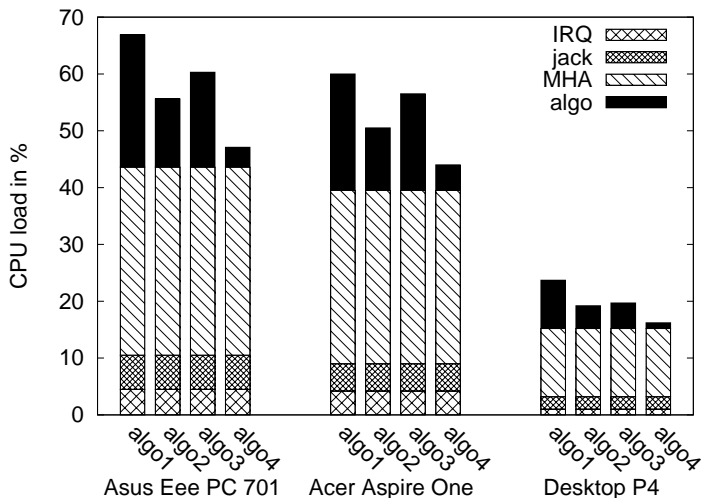


Algorithm implementation in
Matlab

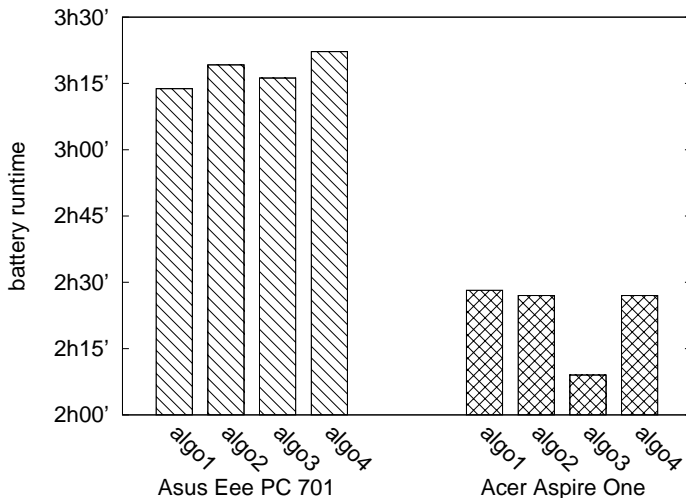
CPU and battttery performance

- ▶ Two portable systems
- ▶ Asus Eee PC 701:
Intel Celeron @ 630 MHz
- ▶ Acer Aspire one:
Intel Atom @ 1.6 GHz
- ▶ Reference Desktop system (Intel P4 @ 3GHz)

Comparison of CPU performance



Comparison of batteries



Delay Constraints

Three sources of delay in RT processing

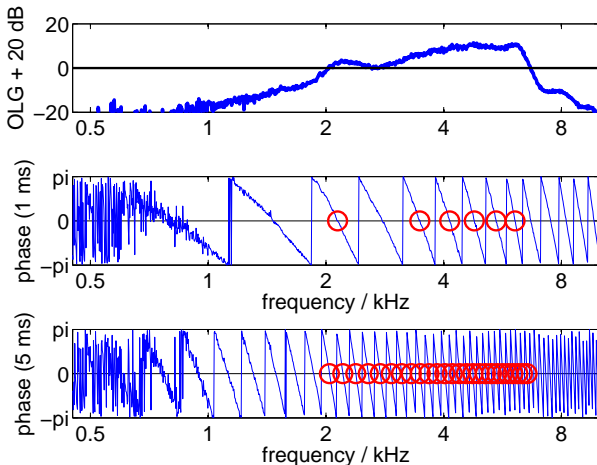
- ▶ Block processing delay:
Fragment size plus processing delay
(i.e., two blocks if input and output is block aligned)
- ▶ Hardware delay:
Anti-aliasing filter (AD and DA, typical 1.5 ms in total)
Data transmission, driver layer ('mystic' delays)
- ▶ Algorithmic delay:
Group delay of filters, overlap-add delay, ...

Why do we aim for lowest delay? \Rightarrow Feedback howling

- ▶ Feedback criterion: Roundtrip-gain is above 0 dB, and phase is multiple of 2π

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Achievable delays

Device	f_s/kHz	fragment size	τ_{sc}/ms	τ_t/ms
Echo Layla 3G	32	32	2.81	4.81
	44.1	64	2.04	4.94
	100	32	0.81	1.77
RME HDSP9652 + Behringer SRC2496	32	64	3.34	7.34
	44.1	64	2.68	5.58
	96	128	1.73	4.40
RME HDSP9652 + Behringer ADA8000	32	64	2.13	6.12
	44.1	64	1.61	4.51
RME HDSP9632 + ADI8QS	44.1	64	1.52	4.42
	96	128	1.03	3.7
OFFIS USB SC-4/2	16	16	6.81	9.81
	44.1	64	4.08	8.44
	96	128	1.97	5.97

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Demonstration now!

Conclusions

- ▶ Hearing aid research is a wide field and requires careful evaluation.
- ▶ Linux Audio provides a valuable environment for hearing aid research.
- ▶ Low delay processing is possible with Linux/ALSA/Jack (but mystic delays remain in the chain of soft- and hardware).

Thank you for your attention!